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Please find below and/or attached an Office communication concerning this application or proceeding.

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DETAILED ACTION

The amendment filed 09/18/2009 has been entered.

- **Claims 1-22 and 24-30 are pending.**
- **Claim 23 is previously cancelled.**
- **Claims 1-5, 8-10, 12-20, 22 and 24-28 stand rejected.**
- **Claims 6, 7, 11, 21, 29 and 30 are objected to.**

Claim Rejections - 35 USC § 103

1. The following is a quotation of 35 U.S.C. 103(a) which forms the basis for all obviousness rejections set forth in this Office action:

(a) A patent may not be obtained though the invention is not identically disclosed or described as set forth in section 102 of this title, if the differences between the subject matter sought to be patented and the prior art are such that the subject matter as a whole would have been obvious at the time the invention was made to a person having ordinary skill in the art to which said subject matter pertains. Patentability shall not be negated by the manner in which the invention was made.

2. The factual inquiries set forth in *Graham v. John Deere Co.*, 383 U.S. 1, 148 USPQ 459 (1966), that are applied for establishing a background for determining obviousness under 35 U.S.C. 103(a) are summarized as follows:

1. Determining the scope and contents of the prior art.
2. Ascertaining the differences between the prior art and the claims at issue.
3. Resolving the level of ordinary skill in the pertinent art.
4. Considering objective evidence present in the application indicating obviousness or nonobviousness.

Claims 1, 2, 8, 9, 15, 16, 19, 24 and 25 are rejected under 35 U.S.C. 103(a) as being unpatentable over Forssell et al. (Document Number: EP 1 006 695 A1) in view of Simard et al. (Patent No.: US 6,940,826 B1) and Powers et al. (Patent No.: US 7,272,660 B1), hereafter respectively referred to as Forssell, Simard, and Powers.

In regard to Claim 1, Forssell teaches in paragraph [0015], lines 47-48, “uplink resource allocation” occurs when the “Mobile Station (MS) requests radio resources”. Forssell also teaches in paragraph [0026], line 15, “Downlink radio resource allocation”. Forssell also teaches in paragraph [0007], from line 58 of page 2 to line 1 of page 3, “packet data transmission between mobile data terminals”, implicitly teaching that transmissions between mobile terminals involve an uplink for a transmitting mobile station and a downlink for a receiving mobile station (communicating in a cellular communications network through a dedicated channel comprising both an uplink and a downlink).

Forssell shows in paragraph [0006], line 33, and Fig. 1a, page 14, “the core network of a cellular system 10” (a core network).

Forssell shows in paragraph [0007], lines 51-55, and Fig. 1b, page 14, the “operational environment comprises one or more subnetwork service areas,” which are interconnected by a backbone network and where each “subnetwork comprises a number of packet data service nodes”, which provide a packet service for mobile data terminals 151 via several base stations 152” (controlling a flow of data packets by at least one of a server function and a server in a core network).

Forssell fails to teach keeping up a dedicated channel downlink from a core network by sending post-speech packets for a time of duration, a server function in a core network transmits post-speech packets to a downlink after receiving a packet indicating an end of speech samples from an uplink, keeping up a dedicated channel after a last speech sample packet is sent downlink from a core network for a time of

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such duration that a new uplink can be established utilizing a downlink from a core network, and a dedicated channel comprising both an uplink and a plurality of downlinks.

Simard teaches in column 6, lines 44-65, and in column 9, lines 22-25, and in column 10, lines 8-33, and in FIGS. 2, 3A, 4, and 5, a talker selection algorithm could transmit empty voice data to terminals within a voice conference when there are no talkers selected in order to maintain continuous packet transmission (a dedicated channel comprising both an uplink and a plurality of downlinks, controlling a flow of data packets by a server function, keeping up a dedicated channel after a last speech packet is sent downlink from a core network by sending post-speech packets for a time of such duration that a new uplink can be established utilizing a downlink from a core network, and wherein a server function transmits post-speech packets to a plurality of downlinks responsive to a packet indicating an end of speech samples from an uplink).

It would have been obvious to one of ordinary skill in the art at the time of the invention to adopt the invention of Simard in the teachings of Forssell since Simard provides a system for voice conferences within packet-based communication networks that provides a reduction in transcoding, latency, and/or required signal processing power (see Simard, column 3, line 65 to column 4, line 15, and column 5, lines 53-59), which can be introduced into the packet-based network of Forssell to allow efficient voice conferencing to users.

Forssell fails to teach controlling a flow of data packets by a server in a core network, keeping up a dedicated channel after a last packet is sent downlink from a

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core network by sending post-speech packets for a time duration, and wherein a server in a core network transmits post-speech packets to a downlink responsive to a packet from an uplink.

Powers teaches in column 5, lines 55-61, and in column 9, lines 54-67, and in FIG. 3A, client applications in a client layer can reside on a Tablet PC, and enabling Windows XP compatible applications to take advantage of various input modes, such as voice-based data, and that a rich client 292 (FIG. 3A) substantially maintains an open TCP-IP connection to active data cache 256 (FIG. 3A), through an execution of a blocking HTTP request in a second open browser window, and a web server periodically send keep-alive messages to a second open browser window, in order to prevent a TCP-IP connection timeout, and through the use of this code, a TCP-IP connection is kept open, allowing data to be sent in a substantially continuous way to rich client 282 (FIG. 3A) without having first to refresh a browser window (controlling a flow of data packets by a server in a core network, keeping up a dedicated channel after a last packet is sent downlink from a core network by sending post-speech packets for a time duration, and wherein a server in a core network transmits post-speech packets to a downlink responsive to a packet from an uplink).

It would have been obvious to one of ordinary skill in the art at the time of the invention to adopt the invention of Powers in the teachings of Forssell since Powers provides a method for optimizing the delivery of data to a device, in which relevant information is received in a timely manner, and in which that data is rendered in a dynamic format (see Powers, column 3, lines 15-36), which can be introduced into the

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system of Forssell to allow data to be dynamically and efficiently updated and provided on user terminals in the system of Forssell.

In regard to Claim 2, Forssell teaches in paragraph [0042], lines 40-41, “the network is informed at the end of an active period, on whether a passive period follows the active period or if the connection can be released” (the server determining when the last speech sample packet is sent).

Forssell teaches in paragraph [0044], lines 53-54, “on an uplink channel, after one mobile station starts to transmit, the other mobile stations may be reallocated to other channels”, and in lines 56-57, “on a downlink channel, after one mobile station starts to transmit, the other mobile stations may be reallocated to other channels” (determining whether a terminal taking part in the session needs a new uplink, establishing said new uplink is established).

Forssell fails to teach “server sending at least one post-speech packet downlink to receiving terminals”.

Simard teaches in column 6, lines 44-65, and in column 9, lines 22-25, and in column 10, lines 8-33, and in FIGS. 2, 3A, 4, and 5, a talker selection algorithm could transmit empty voice data to terminals within a voice conference when there are no talkers selected in order to maintain continuous packet transmission (server sending at least one post-speech packet downlink to receiving terminals).

It would have been obvious to one of ordinary skill in the art at the time of the invention to adopt the invention of Simard in the teachings of Forssell since Simard

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provides a system for voice conferences within packet-based communication networks that provides a reduction in transcoding, latency, and/or required signal processing power (see Simard, column 3, line 65 to column 4, line 15, and column 5, lines 53-59), which can be introduced into the packet-based network of Forssell to allow efficient voice conferencing to users.

In regard to Claims 8 and 9, Forssell shows in paragraph [0006] and in Fig. 1a, page 14, the network of “a cellular radio system” (cellular network).

Forssell teaches in paragraph [0013], lines 38 and 41, that “a Temporary Block Flow (TBF) is created for transferring data packets on a packet data channel” for services that include “voice services” (a last speech sample).

Forssell shows in paragraph [0007], lines 51-55, and Fig. 1b, page 14, the “operational environment comprises one or more subnetwork service areas,” which are interconnected by a backbone network and where each “subnetwork comprises a number of packet data service nodes”, which provide a packet service for mobile data terminals 151 via several base stations 152” (receive a last speech sample packet in an uplink direction in a cellular communications network).

Forssell teaches in paragraph [0042], lines 40-41, that “the network is informed at the end of an active period, on whether a passive period follows the active period”, and, in paragraph [0044], lines 55-56, “on a downlink channel, after one mobile station starts to transmit, the other mobile stations may be reallocated to other channels”, showing that a passive period can occur after an active period on an uplink channel and on a

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downlink channel, and a mobile station of the downlink channel can start transmitting on the channel (a server or a processing device configured to prolong the existence of downlinks for a time of such duration that at least one new uplink can be established from a receiving terminal).

Forssell fails to teach a processing device configured to prolong an existence of downlinks by sending post-speech packets for a time of duration, an apparatus is configured to transmit post-speech packets to downlinks after receiving a packet indicating an end of speech samples from an uplink.

Simard teaches in column 6, lines 44-65, and in column 9, lines 22-25, and in column 10, lines 8-33, and in FIGS. 2, 3A, 4, and 5, a talker selection algorithm could transmit empty voice data to terminals within a voice conference when there are no talkers selected in order to maintain continuous packet transmission (a processing device configured to prolong an existence of downlinks by sending post-speech packets for a time of duration, an apparatus is configured to transmit post-speech packets to downlinks after receiving a packet indicating an end of speech samples from an uplink).

It would have been obvious to one of ordinary skill in the art at the time of the invention to adopt the invention of Simard in the teachings of Forssell since Simard provides a system for voice conferences within packet-based communication networks that provides a reduction in transcoding, latency, and/or required signal processing power (see Simard, column 3, line 65 to column 4, line 15, and column 5, lines 53-59), which can be introduced into the packet-based network of Forssell to allow efficient voice conferencing to users.

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Forssell fails to teach a processing device configured to prolong an existence of a downlink by sending packets for a time of duration, an apparatus is configured to transmit post-speech packets to a downlink after receiving a packet indicating an end from an uplink.

Powers teaches in column 5, lines 55-61, and in column 9, lines 54-67, and in FIG. 3A, client applications in a client layer can reside on a Tablet PC, and enabling Windows XP compatible applications to take advantage of various input modes, such as voice-based data, and that a rich client 292 (FIG. 3A) substantially maintains an open TCP-IP connection to active data cache 256 (FIG. 3A), through an execution of a blocking HTTP request in a second open browser window, and a web server periodically send keep-alive messages to a second open browser window, in order to prevent a TCP-IP connection timeout, and through the use of this code, a TCP-IP connection is kept open, allowing data to be sent in a substantially continuous way to rich client 282 (FIG. 3A) without having first to refresh a browser window (a processing device configured to prolong an existence of a downlink by sending packets for a time of duration, an apparatus is configured to transmit post-speech packets to a downlink after receiving a packet indicating an end from an uplink).

It would have been obvious to one of ordinary skill in the art at the time of the invention to adopt the invention of Powers in the teachings of Forssell since Powers provides a method for optimizing the delivery of data to a device, in which relevant information is received in a timely manner, and in which that data is rendered in a dynamic format (see Powers, column 3, lines 15-36), which can be introduced into the

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system of Forssell to allow data to be dynamically and efficiently updated and provided on user terminals in the system of Forssell.

In regard to Claims 15 and 16, Forssell shows in paragraph [0006] and in Fig. 1a, page 14, the network of “a cellular radio system” (cellular network).

Forssell teaches in paragraph [0013], lines 38 and 41, that “a Temporary Block Flow (TBF) is created for transferring data packets on a packet data channel” for services that include “voice services” (a last speech packet).

Forssell shows in paragraph [0007], lines 51-55, and Fig. 1b, page 14, the “operational environment comprises one or more subnetwork service areas,” which are interconnected by a backbone network and where each “subnetwork comprises a number of packet data service nodes”, which provide a packet service for mobile data terminals 151 via several base stations 152” (a server in a cellular communications network comprising a receiver configured to receive a last speech sample packet in an uplink direction).

Forssell teaches in paragraph [0042], lines 40-41, that “the network is informed at the end of an active period, on whether a passive period follows the active period”, and, in paragraph [0044], lines 55-56, “on a downlink channel, after one mobile station starts to transmit, the other mobile stations may be reallocated to other channels”, showing that a passive period can occur after an active period on an uplink channel and on a downlink channel, and a mobile station of the downlink channel can start transmitting on the channel (a server or a processing device configured to prolong the existence of

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downlinks for a time of such duration that at least one new uplink can be established from a receiving terminal).

Forssell fails to teach a processing device configured to prolong an existence of a downlink by sending post-speech packets for a time of duration, an apparatus is configured to transmit post-speech packets to a downlink after receiving a packet indicating an end of speech samples from an uplink, a server or a processing device configured to prolong the existence of downlinks for a time of such duration that at least one new uplink can be established from a receiving terminal, and a plurality of downlinks.

Simard teaches in column 6, lines 44-65, and in column 9, lines 22-25, and in column 10, lines 8-33, and in FIGS. 2, 3A, 4, and 5, a talker selection algorithm could transmit empty voice data to terminals within a voice conference when there are no talkers selected in order to maintain continuous packet transmission (maintaining a dedicated channel between a sending terminal and a plurality of receiving terminals by sending responsive to a last speech packet from a sending terminal, post speech packets to a plurality of receiving terminals for a time of such duration that a new dedicated channel can be established utilizing an earlier dedicated channel).

It would have been obvious to one of ordinary skill in the art at the time of the invention to adopt the invention of Simard in the teachings of Forssell since Simard provides a system for voice conferences within packet-based communication networks that provides a reduction in transcoding, latency, and/or required signal processing power (see Simard, column 3, line 65 to column 4, line 15, and column 5, lines 53-59),

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which can be introduced into the packet-based network of Forssell to allow efficient voice conferencing to users.

Forssell fails to teach controlling a flow of data packets by a server in a core network, keeping up a dedicated channel after a last packet is sent downlink from a core network by sending post-speech packets for a time duration, and wherein a server in a core network transmits post-speech packets to a downlink responsive to a packet from an uplink.

Powers teaches in column 5, lines 55-61, and in column 9, lines 54-67, and in FIG. 3A, client applications in a client layer can reside on a Tablet PC, and enabling Windows XP compatible applications to take advantage of various input modes, such as voice-based data, and that a rich client 292 (FIG. 3A) substantially maintains an open TCP-IP connection to active data cache 256 (FIG. 3A), through an execution of a blocking HTTP request in a second open browser window, and a web server periodically send keep-alive messages to a second open browser window, in order to prevent a TCP-IP connection timeout, and through the use of this code, a TCP-IP connection is kept open, allowing data to be sent in a substantially continuous way to rich client 282 (FIG. 3A) without having first to refresh a browser window (maintaining a dedicated channel between a sending terminal and a receiving terminal by sending responsive to a last packet from a sending terminal, packets to a terminal for a time of such duration that a new dedicated channel can be established utilizing an earlier dedicated channel).

It would have been obvious to one of ordinary skill in the art at the time of the invention to adopt the invention of Powers in the teachings of Forssell since Powers provides a method for optimizing the delivery of data to a device, in which relevant information is received in a timely manner, and in which that data is rendered in a dynamic format (see Powers, column 3, lines 15-36), which can be introduced into the system of Forssell to allow data to be dynamically and efficiently updated and provided on user terminals in the system of Forssell.

In regard to Claim 19, Forssell teaches in paragraph [0062], lines 28-30, and FIG. 5, page 16, "When the CV' value is set to "0" the network interprets it so that the first mobile station has no more RLC data blocks to be transmitted at the time and the network may therefore give the next N uplink transmit permissions to some other mobile station/stations", and, in lines 34-35, "If the mobile station does not have data to be transmitted, to the network at the time, the mobile station may omit the uplink transmit permission or it may transmit a Packet Dummy Control Block or a signalling message" (an element for sending post-speech packets is a terminal ending its transmission).

In regard to Claim 24, Forssell teaches in paragraph [0015], lines 47-48, "uplink resource allocation" occurs when the "Mobile Station (MS) requests radio resources". Forssell also teaches in paragraph [0026], line 15, "Downlink radio resource allocation". Forssell also teaches in paragraph [0007], from line 58 of page 2 to line 1 of page 3, "packet data transmission between mobile data terminals", implicitly teaching that

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transmissions between mobile terminals involve an uplink for a transmitting mobile station and a downlink for a receiving mobile station (communicating in a cellular communications network through a dedicated channel comprising both an uplink and a downlink).

Forssell shows in paragraph [0006], line 33, and Fig. 1a, page 14, “the core network of a cellular system 10” (a core network).

Forssell shows in paragraph [0007], lines 51-55, and Fig. 1b, page 14, the “operational environment comprises one or more subnetwork service areas,” which are interconnected by a backbone network and where each “subnetwork comprises a number of packet data service nodes”, which provide a packet service for mobile data terminals 151 via several base stations 152” (controlling a flow of data packets by at least one of a server function and a server in a core network).

Forssell further teaches in paragraph [0086], lines 40-42, 47-49, “the processing of information in a telecommunication device takes place in an arrangement of processing capacity in the form of microprocessor(s) and memory in the form of memory circuits. Such arrangements are known as such from the technology of mobile stations and fixed network elements”, and “On the network side, the features according to the invention can be implemented e.g. in the Packet Control Unit PCU”, where “The packet control unit may be located e.g. in the ... Serving GPRS Support Node SGSN” (a computer readable medium encoded with a computer program executable to perform actions).

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Forssell fails to teach keeping up a dedicated channel downlink from a core network by sending post-speech packets for a time of duration, a server function in a core network transmits post-speech packets to a downlink after receiving a packet indicating an end of speech samples from an uplink, keeping up a dedicated channel after a last speech sample packet is sent downlink from a core network for a time of such duration that a new uplink can be established utilizing a downlink from a core network, and a dedicated channel comprising both an uplink and a plurality of downlinks.

Simard teaches in column 6, lines 44-65, and in column 9, lines 22-25, and in column 10, lines 8-33, and in FIGS. 2, 3A, 4, and 5, a talker selection algorithm could transmit empty voice data to terminals within a voice conference when there are no talkers selected in order to maintain continuous packet transmission (a dedicated channel comprising both an uplink and a plurality of downlinks, controlling a flow of data packets by a server function, keeping up a dedicated channel after a last speech packet is sent downlink from a core network by sending post-speech packets for a time of such duration that a new uplink can be established utilizing a downlink from a core network, and wherein a server function transmits post-speech packets to a plurality of downlinks responsive to a packet indicating an end of speech samples from an uplink).

It would have been obvious to one of ordinary skill in the art at the time of the invention to adopt the invention of Simard in the teachings of Forssell since Simard provides a system for voice conferences within packet-based communication networks that provides a reduction in transcoding, latency, and/or required signal processing

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power (see Simard, column 3, line 65 to column 4, line 15, and column 5, lines 53-59), which can be introduced into the packet-based network of Forssell to allow efficient voice conferencing to users.

Forssell fails to teach controlling a flow of data packets by a server in a core network, keeping up a dedicated channel after a last packet is sent downlink from a core network by sending post-speech packets for a time duration, and wherein a server in a core network transmits post-speech packets to a downlink responsive to a packet from an uplink.

Powers teaches in column 5, lines 55-61, and in column 9, lines 54-67, and in FIG. 3A, client applications in a client layer can reside on a Tablet PC, and enabling Windows XP compatible applications to take advantage of various input modes, such as voice-based data, and that a rich client 292 (FIG. 3A) substantially maintains an open TCP-IP connection to active data cache 256 (FIG. 3A), through an execution of a blocking HTTP request in a second open browser window, and a web server periodically send keep-alive messages to a second open browser window, in order to prevent a TCP-IP connection timeout, and through the use of this code, a TCP-IP connection is kept open, allowing data to be sent in a substantially continuous way to rich client 282 (FIG. 3A) without having first to refresh a browser window (controlling a flow of data packets by a server in a core network, keeping up a dedicated channel after a last packet is sent downlink from a core network by sending post-speech packets for a time duration, and wherein a server in a core network transmits post-speech packets to a downlink responsive to a packet from an uplink).

It would have been obvious to one of ordinary skill in the art at the time of the invention to adopt the invention of Powers in the teachings of Forssell since Powers provides a method for optimizing the delivery of data to a device, in which relevant information is received in a timely manner, and in which that data is rendered in a dynamic format (see Powers, column 3, lines 15-36), which can be introduced into the system of Forssell to allow data to be dynamically and efficiently updated and provided on user terminals in the system of Forssell.

In regard to Claim 25, Forssell teaches in paragraph [0042], lines 40-41, “the network is informed at the end of an active period, on whether a passive period follows the active period or if the connection can be released” (the server determining when the last speech sample packet is sent).

Forssell teaches in paragraph [0044], lines 53-54, “on an uplink channel, after one mobile station starts to transmit, the other mobile stations may be reallocated to other channels”, and in lines 56-57, “on a downlink channel, after one mobile station starts to transmit, the other mobile stations may be reallocated to other channels” (determining whether a terminal taking part in the session needs a new uplink, establishing said new uplink is established).

Forssell fails to teach sending at least one post-speech packet downlink to receiving terminals.

Simard teaches in column 6, lines 44-65, and in column 9, lines 22-25, and in column 10, lines 8-33, and in FIGS. 2, 3A, 4, and 5, a talker selection algorithm could

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transmit empty voice data to terminals within a voice conference when there are no talkers selected in order to maintain continuous packet transmission (sending at least one post-speech packet downlink to receiving terminals).

It would have been obvious to one of ordinary skill in the art at the time of the invention to adopt the invention of Simard in the teachings of Forssell since Simard provides a system for voice conferences within packet-based communication networks that provides a reduction in transcoding, latency, and/or required signal processing power (see Simard, column 3, line 65 to column 4, line 15, and column 5, lines 53-59), which can be introduced into the packet-based network of Forssell to allow efficient voice conferencing to users.

Claims 3 and 26 are rejected under 35 U.S.C. 103(a) as being unpatentable over Forssell in view of Simard, Powers, and further in view of Upp et al. (Pub. No.: US 2004/0002351 A1), hereafter referred to as Upp.

In regard to Claim 3, Forssell teaches in paragraph [0034], pages 53-54, that the “network sets the FBI field to ‘1’ when it has no more RLC data blocks to send to the mobile station” (receiving terminal receiving the last speech sample packet).

Forssell fails to teach that the receiving terminal signals the user.

Upp teaches in paragraph [0003], “mobile communication device, which then alerts the user that the channel is open and the user may commence speaking” (receiving terminal signals the user of the terminal).

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It would have been obvious to one skilled in the art at the time of the invention to adopt the method and system for patching dispatch calling parties together and alerting users of Upp with the real time data network of Forssell since it will allow the network to efficiently form and connect talk groups for subscribers.

In regard to Claim 26, Forssell teaches in paragraph [0034], pages 53-54, that the “network sets the FBI field to ‘1’ when it has no more RLC data blocks to send to the mobile station” (receiving terminal receiving the last speech sample packet).

Forssell further teaches in paragraph [0086], lines 40-42,47-49, “the processing of information in a telecommunication device takes place in an arrangement of processing capacity in the form of microprocessor(s) and memory in the form of memory circuits. Such arrangements are known as such from the technology of mobile stations and fixed network elements”, and “On the network side, the features according to the invention can be implemented e.g. in the Packet Control Unit PCU”, where “The packet control unit may be located e.g. in the ... Serving GPRS Support Node SGSN” (a computer readable medium encoded with a computer program executable to perform actions).

Forssell fails to teach that the receiving terminal signals the user.

Upp teaches in paragraph [0003], “mobile communication device, which then alerts the user that the channel is open and the user may commence speaking” (receiving terminal signals the user of the terminal).

It would have been obvious to one skilled in the art at the time of the invention to adopt the method and system for patching dispatch calling parties together and alerting users of Upp with the real time data network of Forssell since it will allow the network to efficiently form and connect talk groups for subscribers.

Claims 4, 5, 10, 20, 27 and 28 are rejected under 35 U.S.C. 103(a) as being unpatentable over Forssell in view of Simard, Powers, and further in view of Lechleider (Patent Number: 6,058,109) and Rinchiuso et al. (Pub. No.: US 2004/0196861 A1), respectively referred to as Lechleider and Rinchiuso.

In regard to Claim 4, Forssell in view of Simard and Powers as applied to Claim 2 teach all those limitations.

Forssell fails to teach a number of post-speech packets to send and intervals in which to send post-speech packets.

Lechleider teaches in lines 34-36, column 6, of a system that “transmits at a rate of 2 packets per second” (post-speech packets are sent...at intervals of 500 ms). Lechleider teaches in lines 35-36, column 6, where a “uniform transmitter packet buffer 250 is 10 packets long”, and Lechleider also teaches in lines 48-49, column 6, where a “packet buffer 330 initially waits until 10 packets have been stored before it initiates transmission”, providing a situation where only 10 packets may be transmitted (substantively the same as “packets are sent...10 times” in the instant invention”). As indicated in line 28, column 6, these are illustrative examples, and Lechleider does not

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exclude that the buffers' operation could involve less than 10 packets (substantively the same as "packets are sent downlink 5 to 10 times" in the instant invention).

It would have been obvious to one skilled in the art at the time of the invention to adopt the system of data transmission during link termination delays of Lechleider into the real time data network of Forssell since it would aid in maximizing the total data transmitted during the active period of a channel.

Rinchuso teaches in paragraph [0031], "the delay period (X) is varied based on the data transmission rate. More particularly, as the data rate increases, the delay will increase proportionally. In the preferred embodiment of the present invention a delay of 200 msec is used for average data rates of 19 KBPS. The delay period is increased linearly to 500 msec for data rates of 100 KBPS. Varying the delay period in proportion to the data transmission rate" (at intervals of 500 ms at most).

It would have been obvious to one skilled in the art at the time of the invention to adopt the channel dropping delay based on data rate system of Rinchuso into the real time data network of Forssell since it "can cut down on the bouncing effect, while minimizing the time period a remote unit needlessly holds" a channel (see paragraph [0031] of Rinchuso).

In regard to Claim 5, Forssell teaches in paragraph [0043], lines 49-51, "The network may use a timer function for determining whether a passive period follows the active period or if the connection can be released." "...when a predetermined time of

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inactive data transfer has passed, the TBF is released” (after the last post-speech packet the downlink used is released after a delay specific to the cellular network).

In regard to Claim 10, Forssell in view of Simard and Powers as applied to Claim 9 teach all those limitations.

Forssell fails to teach a number of post-speech packets to send and intervals in which to send post-speech packets.

Lechleider teaches in lines 34-36, column 6, of a system that “transmits at a rate of 2 packets per second” (post-speech packets are sent...at intervals of 500 ms).

Lechleider teaches in lines 35-36, column 6, where a “uniform transmitter packet buffer 250 is 10 packets long”, and Lechleider also teaches in lines 48-49, column 6, where a “packet buffer 330 initially waits until 10 packets have been stored before it initiates transmission”, providing a situation where only 10 packets may be transmitted (packets are sent 10 times). As indicated in line 28, column 6, these are illustrative examples, and Lechleider does not exclude that the buffers’ operation could involve less than 10 packets (substantively the same as “packets are sent downlink 5 to 10 times” in the instant invention).

It would have been obvious to one skilled in the art at the time of the invention to adopt the system of data transmission during link termination delays of Lechleider into the real time data network of Forssell since it would aid in maximizing the total data transmitted during the active period of a channel.

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Rinchuso teaches in paragraph [0031], “the delay period (X) is varied based on the data transmission rate. More particularly, as the data rate increases, the delay will increase proportionally. In the preferred embodiment of the present invention a delay of 200 msec is used for average data rates of 19 KBPS. The delay period is increased linearly to 500 msec for data rates of 100 KBPS. Varying the delay period in proportion to the data transmission rate” (at intervals of 500 ms at most).

It would have been obvious to one skilled in the art at the time of the invention to adopt the channel dropping delay based on data rate system of Rinchuso into the real time data network of Forssell since it “can cut down on the bouncing effect, while minimizing the time period a remote unit needlessly holds” a channel (see paragraph [0031] of Rinchuso).

In regard to Claim 20, Forssell in view of Simard and Powers as applied to Claim 16 teach those limitations.

Forssell fails to teach a number of post-speech packets to send and intervals in which to send post-speech packets.

Lechleider teaches in lines 34-36, column 6, of a system that “transmits at a rate of 2 packets per second” (post-speech packets are sent...at intervals of 500 ms).

Lechleider teaches in lines 35-36, column 6, where a “uniform transmitter packet buffer 250 is 10 packets long”, and Lechleider also teaches in lines 48-49, column 6, where a “packet buffer 330 initially waits until 10 packets have been stored before it initiates transmission”, providing a situation where only 10 packets may be transmitted

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(substantively the same as “packets are sent...10 times” in the instant invention”). As indicated in line 28, column 6, these are illustrative examples, and Lechleider does not exclude that the buffers’ operation could involve less than 10 packets (packets are sent downlink 5 to 10 times).

It would have been obvious to one skilled in the art at the time of the invention to adopt the system of data transmission during link termination delays of Lechleider into the real time data network of Forssell since it would aid in maximizing the total data transmitted during the active period of a channel.

Rinchuso teaches in paragraph [0031], “the delay period (X) is varied based on the data transmission rate. More particularly, as the data rate increases, the delay will increase proportionally. In the preferred embodiment of the present invention a delay of 200 msec is used for average data rates of 19 KBPS. The delay period is increased linearly to 500 msec for data rates of 100 KBPS. Varying the delay period in proportion to the data transmission rate” (at intervals of 500 ms at most).

It would have been obvious to one skilled in the art at the time of the invention to adopt the channel dropping delay based on data rate system of Rinchuso into the real time data network of Forssell since it “can cut down on the bouncing effect, while minimizing the time period a remote unit needlessly holds” a channel (see paragraph [0031] of Rinchuso).

In regard to Claim 27, Forssell in view of Simard and Powers as applied to Claim 25 teach all those limitations.

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Forssell fails to teach a number of post-speech packets to send and intervals in which to send post-speech packets.

Lechleider teaches in lines 34-36, column 6, of a system that “transmits at a rate of 2 packets per second” (substantively the same as “post-speech packets are sent...at intervals of 500 ms” in the instant invention). Lechleider teaches in lines 35-36, column 6, where a “uniform transmitter packet buffer 250 is 10 packets long”, and Lechleider also teaches in lines 48-49, column 6, where a “packet buffer 330 initially waits until 10 packets have been stored before it initiates transmission”, providing a situation where only 10 packets may be transmitted (substantively the same as “packets are sent...10 times” in the instant invention”). As indicated in line 28, column 6, these are illustrative examples, and Lechleider does not exclude that the buffers’ operation could involve less than 10 packets (substantively the same as “packets are sent downlink 5 to 10 times” in the instant invention).

It would have been obvious to one skilled in the art at the time of the invention to adopt the system of data transmission during link termination delays of Lechleider into the real time data network of Forssell since it would aid in maximizing the total data transmitted during the active period of a channel.

Rinchiuso teaches in paragraph [0031], “the delay period (X) is varied based on the data transmission rate. More particularly, as the data rate increases, the delay will increase proportionally. In the preferred embodiment of the present invention a delay of 200 msec is used for average data rates of 19 KBPS. The delay period is increased linearly to 500 msec for data rates of 100 KBPS. Varying the delay period in proportion

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to the data transmission rate” (substantively the same as “at intervals of 500 ms at most” in the instant invention).

It would have been obvious to one skilled in the art at the time of the invention to adopt the channel dropping delay based on data rate system of Rinchiuso into the real time data network of Forssell since it “can cut down on the bouncing effect, while minimizing the time period a remote unit needlessly holds” a channel (see paragraph [0031] of Rinchiuso).

In regard to Claim 28, Forssell teaches in paragraph [0043], lines 49-51, “The network may use a timer function for determining whether a passive period follows the active period or if the connection can be released.” “...when a predetermined time of inactive data transfer has passed, the TBF is released” (substantively the same as “after the last post-speech packet the downlink used is released after a delay specific to the cellular network” in the instant invention).

Claim 12 is rejected under 35 U.S.C. 103(a) as being unpatentable over Forssell in view of Simard.

In regard to Claim 12, Forssell shows in paragraph [0084], and FIG. 10, page 20, shows a “block diagram of a mobile station 100”, where a control unit (item 103) is substantively the same as the control unit of applicant, a RR-receiver, A/D-converter (item 111) is substantively the same as the receiver RX of applicant, a memory (item 104) is substantively the same as the memory of applicant, a modulator, RF-transmitter

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(item 123) is substantively the same as the transmitter TX of applicant, and a keyboard (item 131) and a display (item 132) are substantively the same as the user interface UI of applicant (cellular terminal, comprising a control unit).

Forssell fails to teach a terminal recognizing post-speech packets which are transferable on a packet data channel.

Simard teaches in column 6, lines 44-65, and in column 9, lines 22-25, and in column 10, lines 8-33, and in FIGS. 2, 3A, 4, and 5, a talker selection algorithm could transmit empty voice data to terminals within a voice conference when there are no talkers selected in order to maintain continuous packet transmission (a terminal recognizing post-speech packets which are transferable on a packet data channel).

It would have been obvious to one of ordinary skill in the art at the time of the invention to adopt the invention of Simard in the teachings of Forssell since Simard provides a system for voice conferences within packet-based communication networks that provides a reduction in transcoding, latency, and/or required signal processing power (see Simard, column 3, line 65 to column 4, line 15, and column 5, lines 53-59), which can be introduced into the packet-based network of Forssell to allow efficient voice conferencing to users.

Claim 13 is rejected under 35 U.S.C. 103(a) as being unpatentable over Forssell in view of Simard, and further in view of Upp.

In regard to Claim 13, Forssell teaches in paragraph [0034], pages 53-54, that the “network sets the FBI field to ‘1’ when it has no more RLC data blocks to send to the mobile station” (after receiving a last speech sample packet).

Forssell fails to teach that the receiving terminal signals the user.

Upp teaches in paragraph [0003], “mobile communication device, which then alerts the user that the channel is open and the user may commence speaking” (substantively the same as “a control unit further configures to perform signaling” in the instant invention).

It would have been obvious to one skilled in the art at the time of the invention to adopt the method and system for patching dispatch calling parties together and alerting users of Upp with the real time data network of Forssell since it will allow the network to efficiently form and connect talk groups for subscribers.

Claim 14 is rejected under 35 U.S.C. 103(a) as being unpatentable over Forssell in view of Simard, and further in view of Kajizaki.

In regard to Claim 14, Forssell in view of Schieder as applied to Claim 12 teach all those limitations.

Forssell fails to teach appending packets together.

Kajizaki teaches in the abstract, “When a routing processing unit detects the transmission of a ... number of packets addressed to the same destination ... A combining unit assembles a combined packet” (where the received post-speech packets are appended to speech sample packets).

It would have been obvious to one skilled in the art at the time of the invention to adopt the packet combining of Kajizaki into the real time data network of Forssell since packets below a certain size can result in unacceptable overhead and inefficient link performance.

Claims 17 and 18 are rejected under 35 U.S.C. 103(a) as being unpatentable over Forssell in view of Simard, Powers, and further in view of Schieder.

In regard to Claim 17, as discussed in the rejection of Claim 15, Forssell in view of Cromer teaches a network post-speech packets.

Forssell fails to teach that non-speech packets are sent from a server operating in the network.

Schieder shows in paragraph [0035], lines 51-54, and FIG. 5a, page 23, item ST5a2, that the network side will transmit a non-data message to the subscriber terminal after the subscriber terminal has finished sending data packets. Schieder also shows in paragraph [0087], lines 22-26, and FIG. 10, page 29, item ST102, that the network side NS will transmit a non-data packet after the network side NS has finished sending data packets. In both cases, it is not specified which network side (NS) element of FIG. 1, page 18, sends the non-data message and packet. As a result, Schieder implicitly teaches that any one of the network side (NS) element could be the origination of the non-data message or packet. Schieder teaches in paragraph [0006], lines 50-55, and FIG. 1, that a network side (NS) element is a SGSN, where a "node SGSN (SGSN: Serving GPRS Support Node) is provided which is interfaced via

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interfaces Gb, Gs, Gr with the base station controller BSC, the mobile switching centre MSC and the home location register HLR. Via the SGSN node an IP backbone network can be accessible in the conventional mobile communication network.” (Substantively the same as “an element for sending post-speech packets is a server operating in the network” in the instant invention).

It would have been obvious to one skilled in the art at the time of the invention to combine aspects of the real time data network of Forssell with the network controller and communication system of Schieder since, in the network of Forssell, “the physical connection of a packet radio service is kept reserved during the passive periods of a session but the same physical resources can still be shared between multiple users” (see abstract of Forssell), and the uplink acknowledgement/negative acknowledgement message of the network of Scheider can be used in the system of Forssell so that a network side can acknowledge to a transmitting mobile station that the last data packet is received in the uplink channel and can also contain information related to channel and network maintenance or information informing the mobile station that the network side has data packets addressed to the mobile station.

In regard to Claim 18, Forssell in view of Simard, Powers, and Schieder as applied to Claim 17 teach all those limitations.

Forssell fails to teach a router server.

Schieder shows in FIG. 2, page 19, that the SGSN (see also FIG. 1, page 18, item SGSN) operates with the Layer 3, IP-based protocols SMDCP and GTP, teaching

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that the SGSN provides routing functions (substantively the same as “the server sending post-speech packets is a router server” in the instant invention).

It would have been obvious to one skilled in the art at the time of the invention to combine aspects of the real time data network of Forssell with the network controller and communication system of Schieder since, in the network of Forssell, “the physical connection of a packet radio service is kept reserved during the passive periods of a session but the same physical resources can still be shared between multiple users” (see abstract of Forssell), and the uplink acknowledgement/negative acknowledgement message of the network of Scheider can be used in the system of Forssell so that a network side can acknowledge to a transmitting mobile station that the last data packet is received in the uplink channel and can also contain information related to channel and network maintenance or information informing the mobile station that the network side has data packets addressed to the mobile station.

Claim 22 is rejected under 35 U.S.C. 103(a) as being unpatentable over Forssell in view of Simard.

In regard to Claim 22, Forssell teaches in paragraph [0042], lines 40-41, that “the network is informed at the end of an active period, on whether a passive period follows the active period”, and, in paragraph [0044], lines 55-56, “on a downlink channel, after one mobile station starts to transmit, the other mobile stations may be reallocated to other channels”, showing that a passive period can occur after an active period on an uplink channel and on a downlink channel, and a mobile station of the downlink channel

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can start transmitting on the channel (determining whether a receiving terminal taking part in a session needs a new uplink, and establishing an uplink).

Forssell further teaches in paragraph [0086], lines 40-42,47-49, “the processing of information in a telecommunication device takes place in an arrangement of processing capacity in the form of microprocessor(s) and memory in the form of memory circuits. Such arrangements are known as such from the technology of mobile stations and fixed network elements”, and “On the network side, the features according to the invention can be implemented e.g. in the Packet Control Unit PCU”, where “The packet control unit may be located e.g. in the ... Serving GPRS Support Node SGSN” (a data storage medium encoded with software readable by a data processing device for performing actions).

Forssell fails to teach determining when a last speech sample packet is sent uplink, sending a post-speech packet to a plurality of receiving terminals responsive to a last speech sample.

Simard teaches in column 6, lines 44-65, and in column 9, lines 22-25, and in column 10, lines 8-33, and in FIGS. 2, 3A, 4, and 5, a talker selection algorithm could transmit empty voice data to terminals within a voice conference when there are no talkers selected in order to maintain continuous packet transmission (determining when a last speech sample packet is sent uplink, sending a post-speech packet to a plurality of receiving terminals responsive to a last speech sample).

It would have been obvious to one of ordinary skill in the art at the time of the invention to adopt the invention of Simard in the teachings of Forssell since Simard

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provides a system for voice conferences within packet-based communication networks that provides a reduction in transcoding, latency, and/or required signal processing power (see Simard, column 3, line 65 to column 4, line 15, and column 5, lines 53-59), which can be introduced into the packet-based network of Forssell to allow efficient voice conferencing to users.

Allowable Subject Matter

Claims 6 and 7 are objected to as being dependent upon a rejected base claim, but would be allowable if rewritten in independent form including all of the limitations of the base claim and any intervening claims.

Claim 11 is objected to as being dependent upon a rejected base claim, but would be allowable if rewritten in independent form including all of the limitations of the base claim and any intervening claims.

Claim 21 is objected to as being dependent upon a rejected base claim, but would be allowable if rewritten in independent form including all of the limitations of the base claim and any intervening claims.

Claims 29 and 30 are objected to as being dependent upon a rejected base claim, but would be allowable if rewritten in independent form including all of the limitations of the base claim and any intervening claims.

Response to Arguments

I. Arguments for Claim Objections.

Applicant's arguments, see pages 8-9, filed 09/18/2009, with respect to the Claim Objections have been fully considered and are persuasive. The Claim objections have been withdrawn.

II. Arguments for Claim Rejections under 35 USC § 103.

Applicant's arguments filed 09/18/2009 have been fully considered but they are not persuasive. Applicants submit that Forssell does not address the same problems as those addressed in the application. Examiner respectfully disagrees this is sufficient for the withdrawal of the claim rejections. Although Forssell does not explicitly state an interest in solving the same problem as the application, Forssell provides an operation that is pertinent to solving the problem, where, Forssell teaches in paragraph [0042], lines 40-41, that "the network is informed at the end of an active period, on whether a passive period follows the active period", showing that a passive period can occur after an active period on an uplink channel and on a downlink channel, and, in paragraph [0044], lines 55-56, "on a downlink channel, after one mobile station starts to transmit, the other mobile stations may be reallocated to other channels", a mobile station of the downlink channel can have an opportunity to start transmitting on the channel.

Applicant also submit that a person skilled in the art would not be motivated to modify Forssell to transmit post-speech packets responsive to a packet indicating an

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end of speech samples from the uplink, as in Claim 1. Examiner respectfully disagrees this is sufficient for the withdrawal of the claim rejections. Post-speech packets can be used by the mobile device of Forssell to send updates to the network to inform the network if the channel reservation of the last RCL data block has become invalid because the user decides not to continue communications and the channel is no longer needed as originally anticipated, or if the TBF release indication of the last RCL data block has become invalid because the user decides to continue communications and the channel needs to be persevered before the tearing down process is complete, and this would not conflict with or eliminate the need for the CV, TR, and/or FBI parameters of the MAC header received from the mobile device, and could actually be used in the post-speech packets since it would not require the development of a different header format for use in the post-speech packets.

Applicants also submit that the references cited can not be seen to disclose or suggest at least where independent claim 12 recites in part "at least one of recognize or transmit post-speech packets on a packet data channel responsive to a packet indicating an end of speech samples". Examiner respectfully disagrees this is sufficient for the withdrawal of the claim rejections. As discussed in the rejection of claim 12, Simard teaches in column 6, lines 44-65, and in column 9, lines 22-25, and in column 10, lines 8-33, and in FIGS. 2, 3A, 4, and 5, a talker selection algorithm could transmit empty voice data to terminals within a voice conference when there are no talkers selected in order to maintain continuous packet transmission. As a result, Simard implicitly teaches that the terminals recognize the empty voice data in some fashion.

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Since empty voice packets are still voice packets, and these empty voice packets are being transmitted to the terminals, this implies that the empty voice packets are still being received and processed by the terminals. Although no audio is being outputted from audio output components of the terminals, the empty voice packets are still being received and processed by the terminals. Simard does not explicitly or implicitly teach that the empty voice packets are not being received and processed.

Applicants also submit that the references cited can not be seen to disclose or suggest at least where independent claim 15 recites in part "sending, responsive to a last speech packet from the sending terminal, post speech packets to the plurality of receiving terminals for a time of such duration that a new dedicated channel can be established utilizing said earlier dedicated channel". Examiner respectfully disagrees this is sufficient for the withdrawal of the claim rejections. As discussed in the rejection of Claim 15, Simard teaches in column 6, lines 44-65, and in column 9, lines 22-25, and in column 10, lines 8-33, and in FIGS. 2, 3A, 4, and 5, a talker selection algorithm could transmit empty voice data to terminals within a voice conference when there are no talkers selected in order to maintain continuous packet transmission. As a result, Simard implicitly teaches that, during a voice conference, if a single person stops talking, and this results in no person talking, empty voice packets are transmitted to maintain continuous packet transmission, and this transmission of empty voice packets will cease if a person starts talking, and this person talking can be another person participating in the conference call. This situation of a conference call allows another person, who was previously listening, to start talking, and this is substantively the same

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as maintaining a dedicated channel between a sending terminal and a plurality of receiving terminals by sending responsive to a last speech packet from a sending terminal, post speech packets to a plurality of receiving terminals for a time of such duration that a new dedicated channel can be established utilizing an earlier dedicated channel, when the teachings of Simard is combined with the teachings of Forssell and Powers.

Conclusion

3. **THIS ACTION IS MADE FINAL.** Applicant is reminded of the extension of time policy as set forth in 37 CFR 1.136(a).

A shortened statutory period for reply to this final action is set to expire THREE MONTHS from the mailing date of this action. In the event a first reply is filed within TWO MONTHS of the mailing date of this final action and the advisory action is not mailed until after the end of the THREE-MONTH shortened statutory period, then the shortened statutory period will expire on the date the advisory action is mailed, and any extension fee pursuant to 37 CFR 1.136(a) will be calculated from the mailing date of the advisory action. In no event, however, will the statutory period for reply expire later than SIX MONTHS from the mailing date of this final action.

Any inquiry concerning this communication or earlier communications from the examiner should be directed to JOSHUA SMITH whose telephone number is 571-270-1826. The examiner can normally be reached on Monday-Friday, 10:30am-7pm, EST.

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If attempts to reach the examiner by telephone are unsuccessful, the examiner's supervisor, Chirag Shah can be reached on 571-272-3144. The fax phone number for the organization where this application or proceeding is assigned is 571-273-8300.

Information regarding the status of an application may be obtained from the Patent Application Information Retrieval (PAIR) system. Status information for published applications may be obtained from either Private PAIR or Public PAIR. Status information for unpublished applications is available through Private PAIR only. For more information about the PAIR system, see <http://pair-direct.uspto.gov>. Should you have questions on access to the Private PAIR system, contact the Electronic Business Center (EBC) at 866-217-9197 (toll-free). If you would like assistance from a USPTO Customer Service Representative or access to the automated information system, call 800-786-9199 (IN USA OR CANADA) or 571-272-1000.

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Patent Examiner
12-31-2009

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1-3-10